

## CLAIMS

1. A method of estimating the signal-to-noise ratio of a wanted signal, in particular a digital signal, received by a radiocommunications receiver, characterized in that, to minimize the estimation noise of the signal-to-noise ratio, the signal and the noise are estimated separately and the signal ( $E_p$ ) and the noise ( $N_0$ ) are filtered (36, 44) separately before division (40) of the signal by the noise.
2. A method according to claim 1, characterized in that the filtering (36) of the wanted signal ( $E_p$ ) is different from the filtering (44) of the noise signal ( $N_0$ ).
- ~~3. A method according to claim 1 or claim 2, characterized in that, to filter the noise signal, the statistical distribution of the noise power measurements is observed for a particular period (T) during which a statistically representative number of measurement samples is collected and which is sufficiently short for the noise to remain practically stationary.~~
4. A method according to claim 3, characterized in that the noise level used has a value ( $\mu_{N_0} + \Delta_{N_0}$ ) such that the probability (P) that the noise level exceeds that value is less than a predetermined threshold ( $\epsilon$ ) during the observation period (T).
- ~~5. A method according to claim 3 or claim 4, characterized in that the noise value used is the maximum value over the particular period (T).~~
- ~~6. A method according to claim 3 or claim 4, characterized in that the moments of the distribution are determined.~~
7. A method according to claim 6, characterized in that

the average ( $\mu$ ) and the variance ( $\sigma^2$ ) of the distribution are determined and in that the noise value used is  $\mu + n\sigma$ , where  $\sigma$  is a standard deviation and  $n$  is a number determined according to the predetermined threshold.

5 ~~8. A method according to claim 1 or claim 2,~~  
characterized in that a finite or infinite impulse  
response low-pass filter is used to filter the noise  
signal.

9. A method according to any preceding claim,  
10 characterized in that a finite impulse response filter is  
~~used to filter the wanted signal ( $E_p$ ).~~

10. A method according to claim 9, characterized in that  
the finite impulse response filter is an averaging  
filter.

15 ~~11. A method according to claim 9 or claim 10,~~  
characterized in that the transmitter provides a  
reference signal with a regular period at a particular  
level and the signal-to-noise ratio is estimated from  
that reference signal.

20 ~~12. A method according to any of claims 1 to 8,~~  
characterized in that an infinite impulse response filter  
~~is used to filter the estimate of the wanted signal.~~

13. A method according to claim 12, characterized in  
that a first order auto-regressive filter is used, for  
25 example, as expressed by the equation:

$$\hat{x}_i = (1-a)\tilde{x}_i + a\hat{x}_{i-1}$$

where  $\tilde{x}_i$  represents the instantaneous estimate of the  
wanted signal at time  $i$ ,  $\hat{x}_i$  represents the smoothed  
estimate of the wanted signal at time  $i$  and  $a$  is an  
30 integration coefficient.

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15. An application of the method according to any preceding claim to estimating the signal-to-noise ratio in a telecommunications receiver sending data for ~~controlling the power of a corresponding transmitter.~~

$\mathcal{H}^1(\mathbb{R}^n)$  is the space of functions of bounded variation on  $\mathbb{R}^n$ . The space  $\mathcal{H}^1(\mathbb{R}^n)$  is a Banach space with the norm  $\|u\|_{\mathcal{H}^1(\mathbb{R}^n)} = \|u\|_{L^1(\mathbb{R}^n)} + \|\nabla u\|_{L^1(\mathbb{R}^n)}$ . The space  $\mathcal{H}^1(\mathbb{R}^n)$  is a subspace of  $L^1(\mathbb{R}^n)$ . The space  $\mathcal{H}^1(\mathbb{R}^n)$  is a Banach space with the norm  $\|u\|_{\mathcal{H}^1(\mathbb{R}^n)} = \|u\|_{L^1(\mathbb{R}^n)} + \|\nabla u\|_{L^1(\mathbb{R}^n)}$ . The space  $\mathcal{H}^1(\mathbb{R}^n)$  is a subspace of  $L^1(\mathbb{R}^n)$ .